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Reuven

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## [54] APPARATUS AND METHOD FOR TRANSMISSION OF HIGH SPEED DATA OVER COMMUNICATION CHANNELS

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[51] Int. Cl.<sup>7</sup> ..... **H04B 1/38; H04M 1/00**

[52] U.S. Cl. .... **375/222; 379/410**

[58] Field of Search ..... 375/222, 219, 375/261, 298, 316, 358, 340; 333/14; 379/416, 406, 410, 411; 370/286, 282, 289, 290

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Primary Examiner—Chi H. Pham

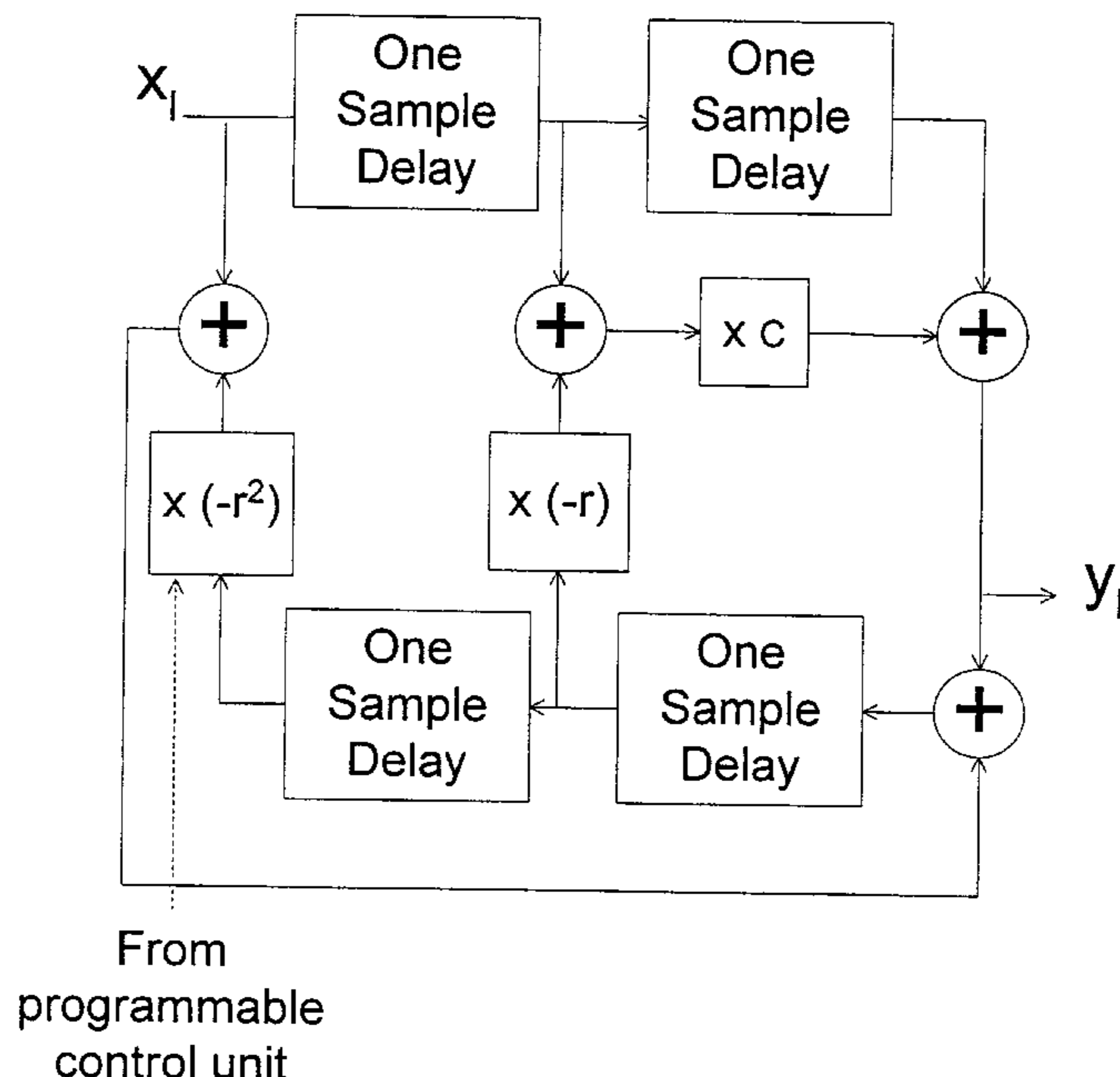
Assistant Examiner—Khai Tran

Attorney, Agent, or Firm—Darby & Darby

### [57] ABSTRACT

This invention discloses an apparatus for transmission of high speed data over communication channels, the apparatus including a modulator operative to modulate an outgoing stream of digital data, thereby to generate an outgoing signal, and a demodulator operative to demodulate an incoming signal, thereby to generate an incoming stream of digital data, wherein the modulator comprises a band suppressor for suppressing portions of the outgoing signal which have specified frequencies.

10 Claims, 10 Drawing Sheets



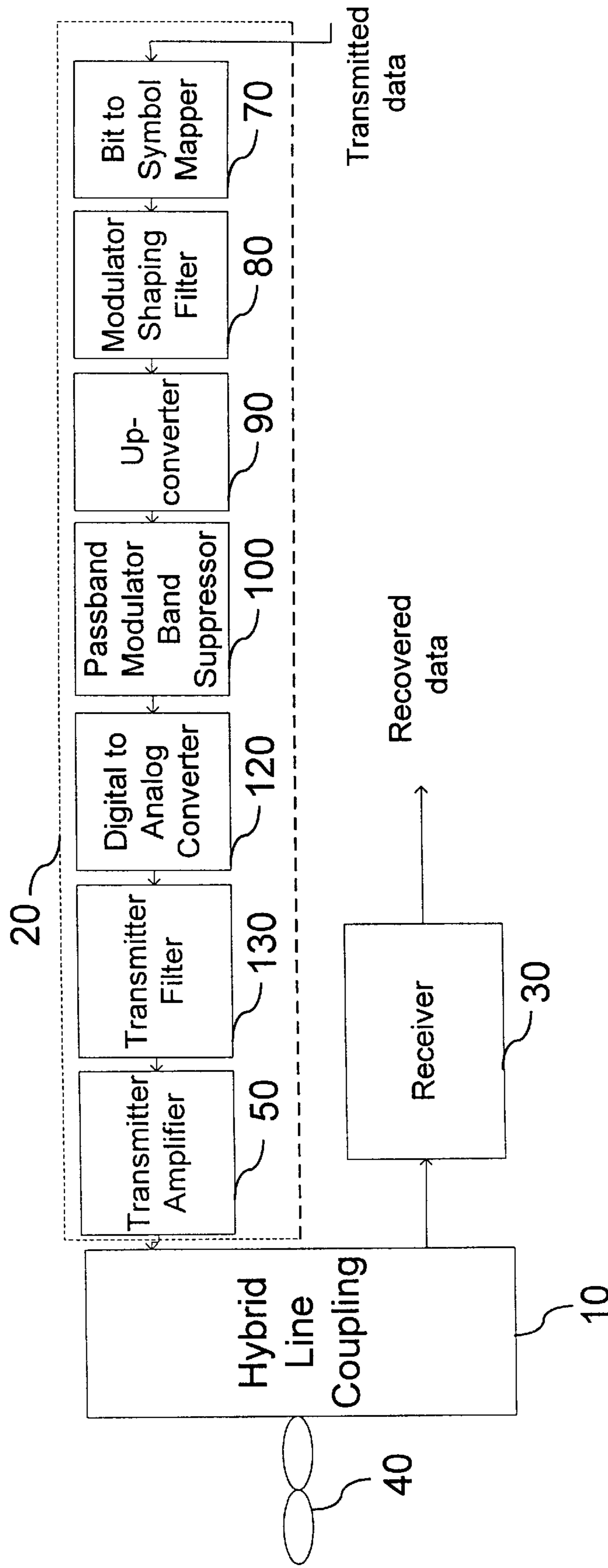


Fig. 1A

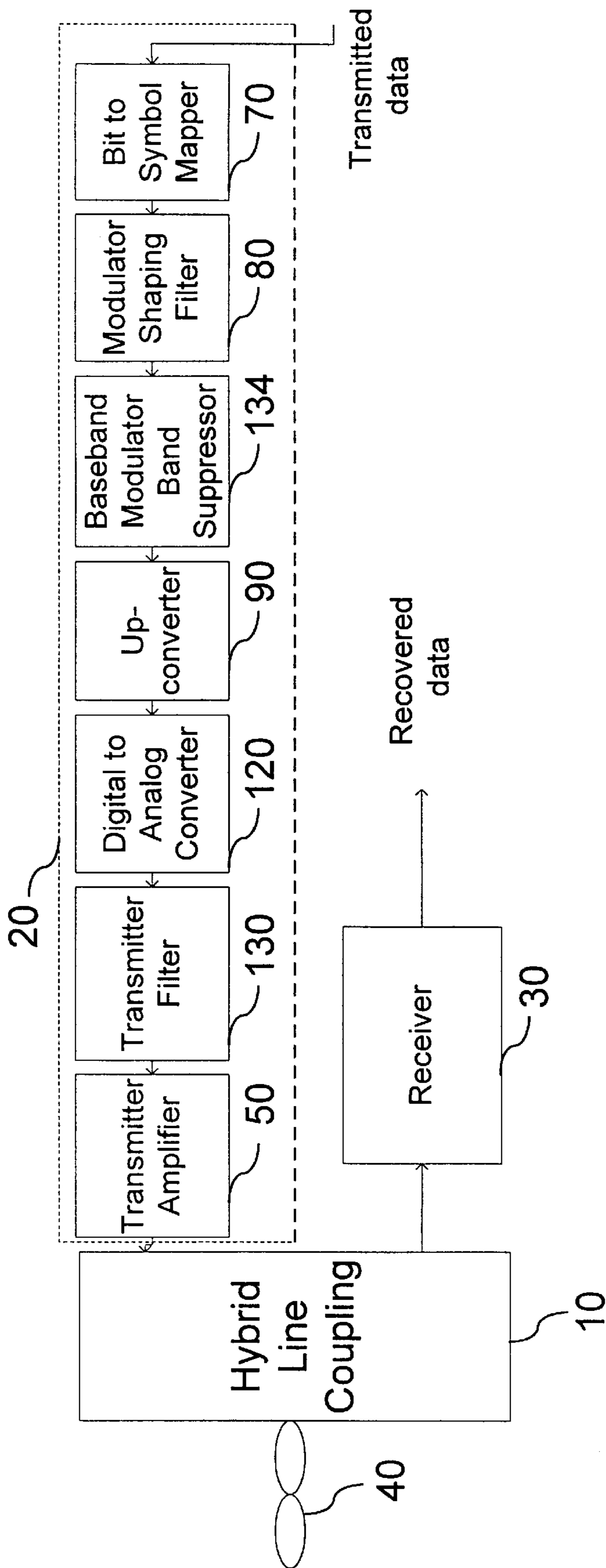


Fig. 1B

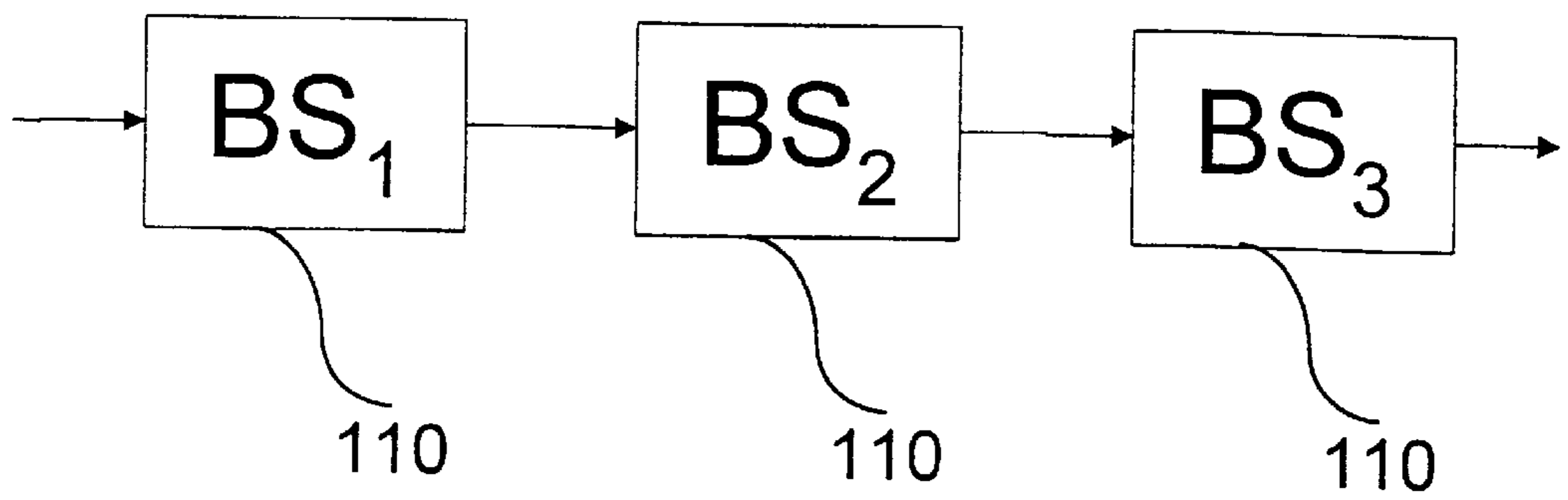


Fig. 2

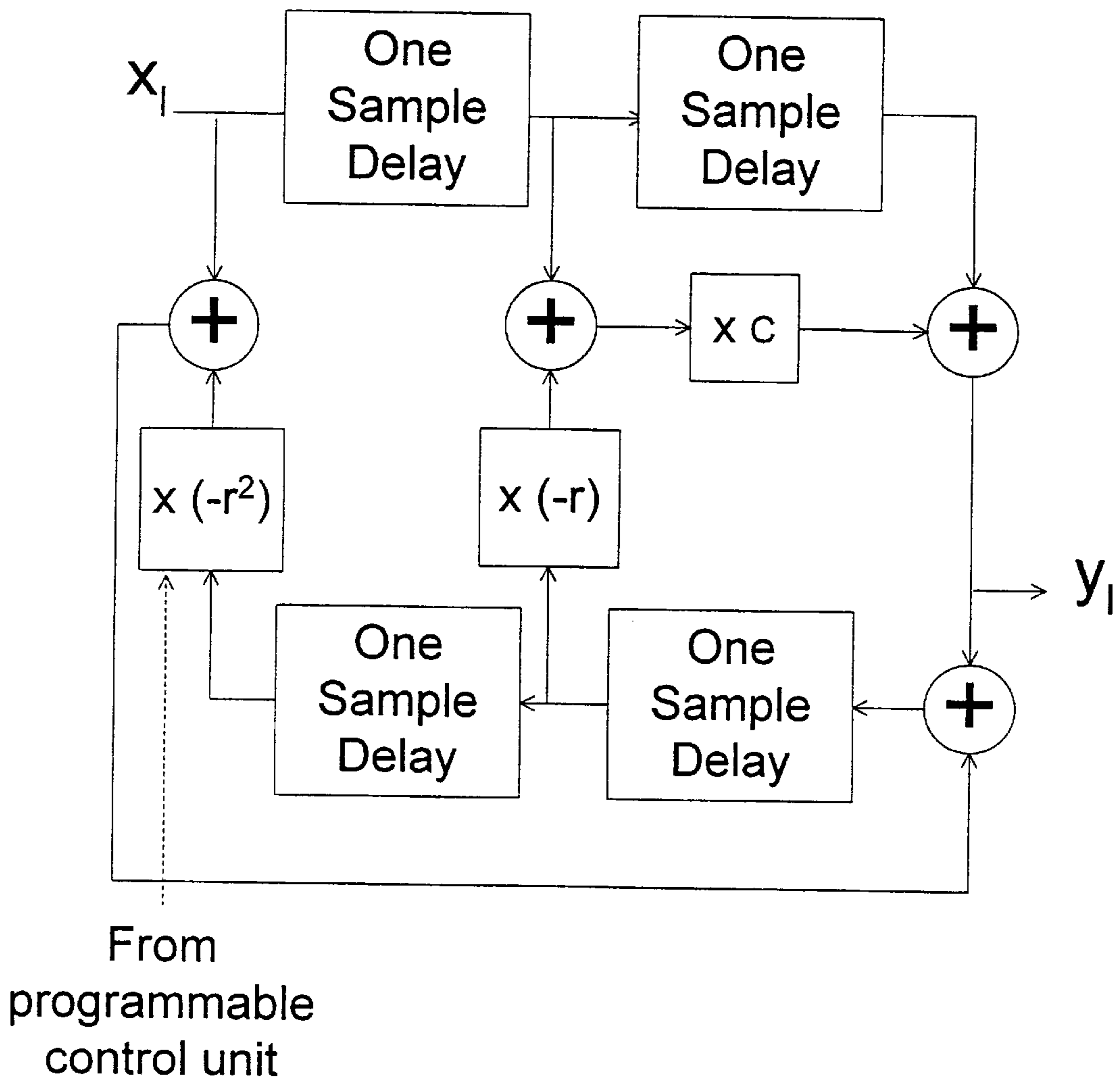


Fig. 3

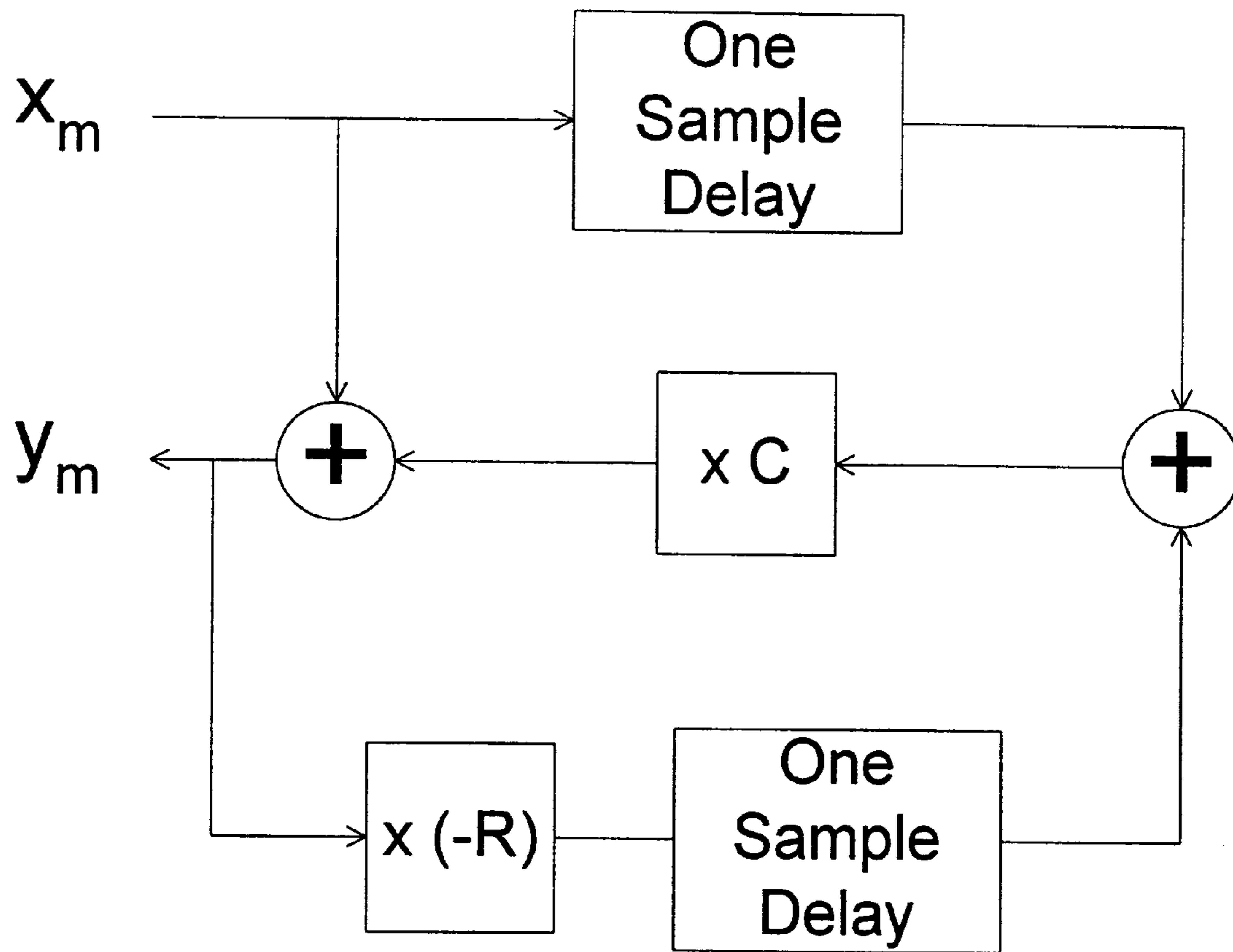


Fig. 4

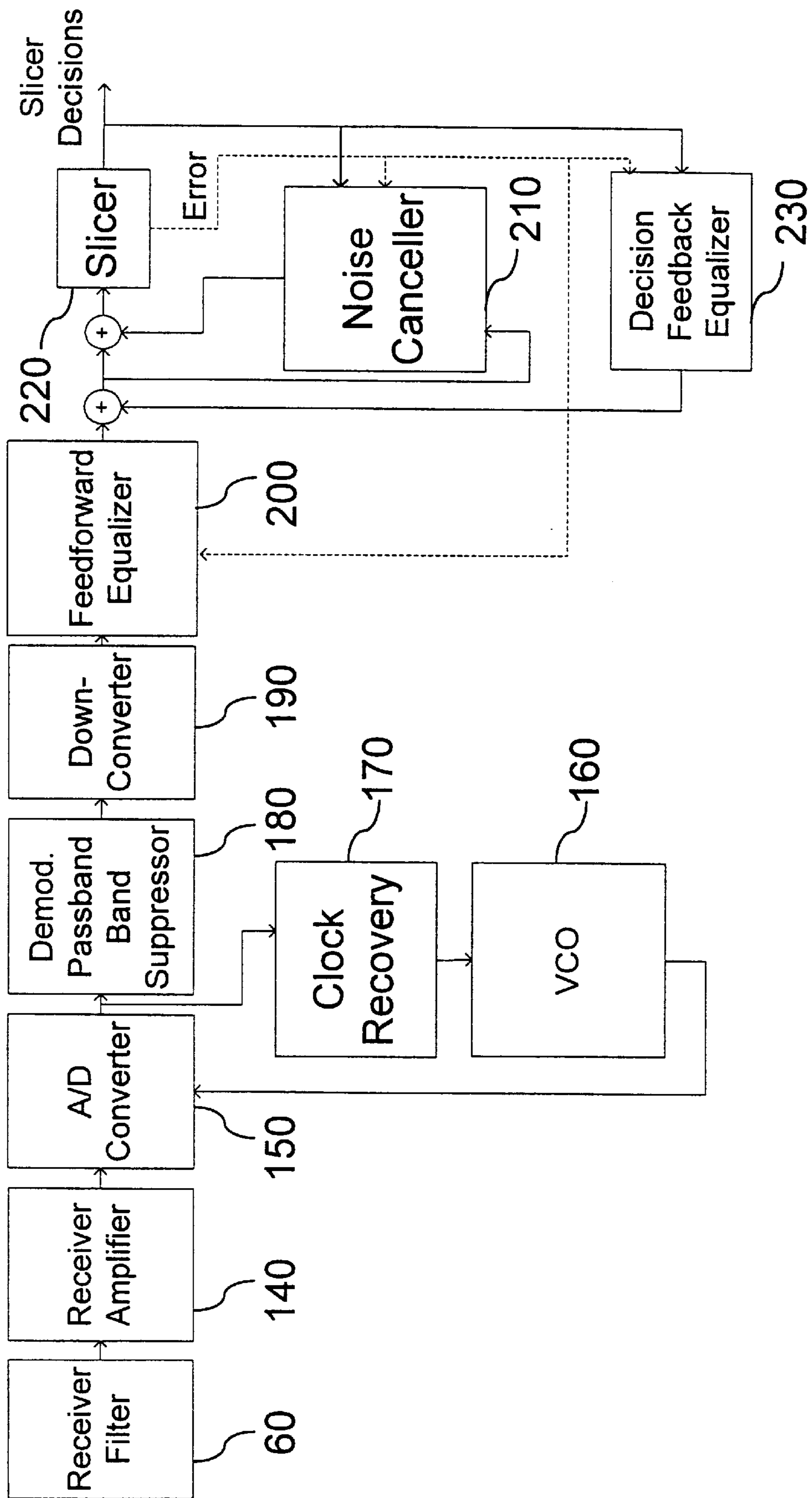


Fig. 5A



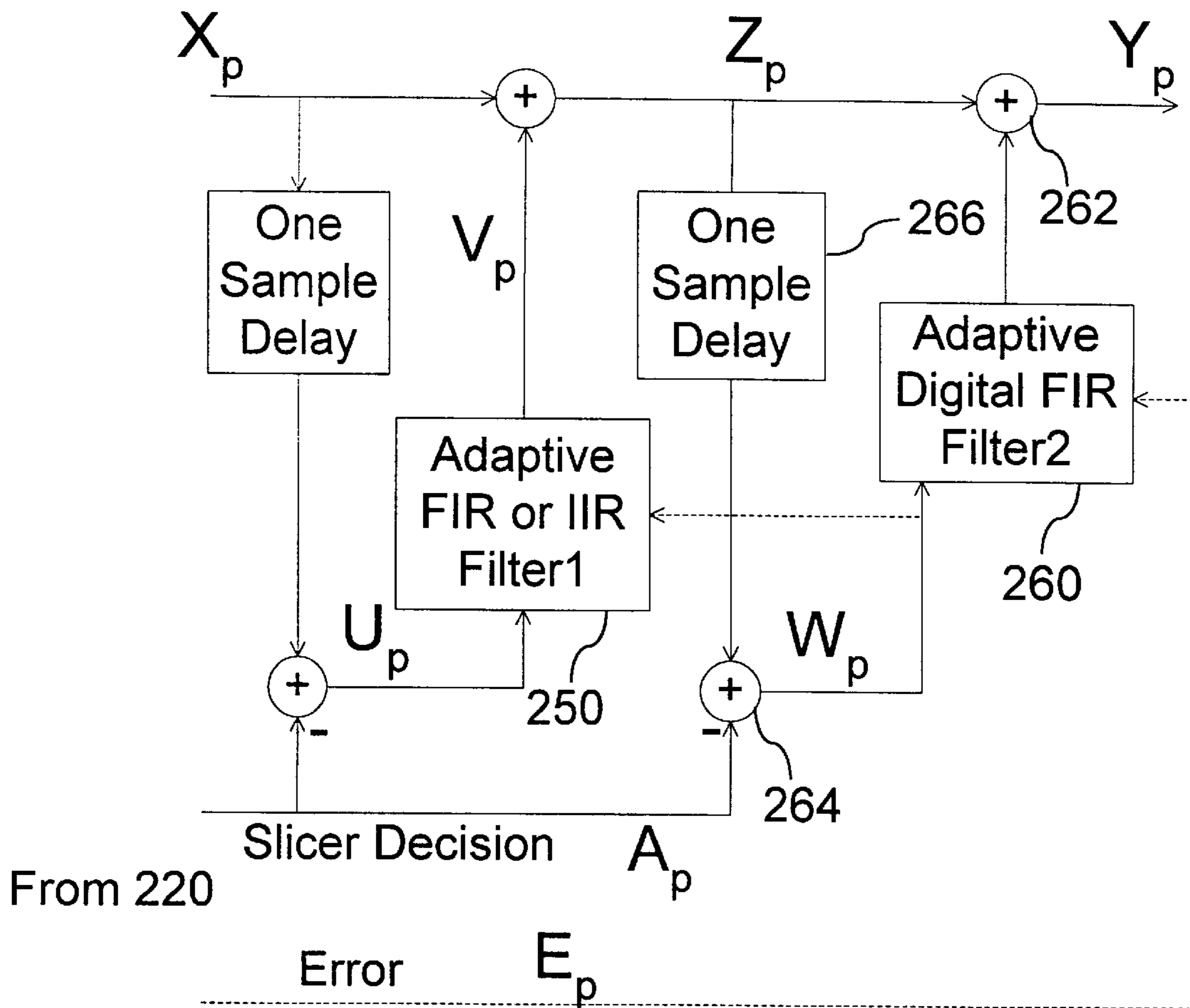


Fig. 6



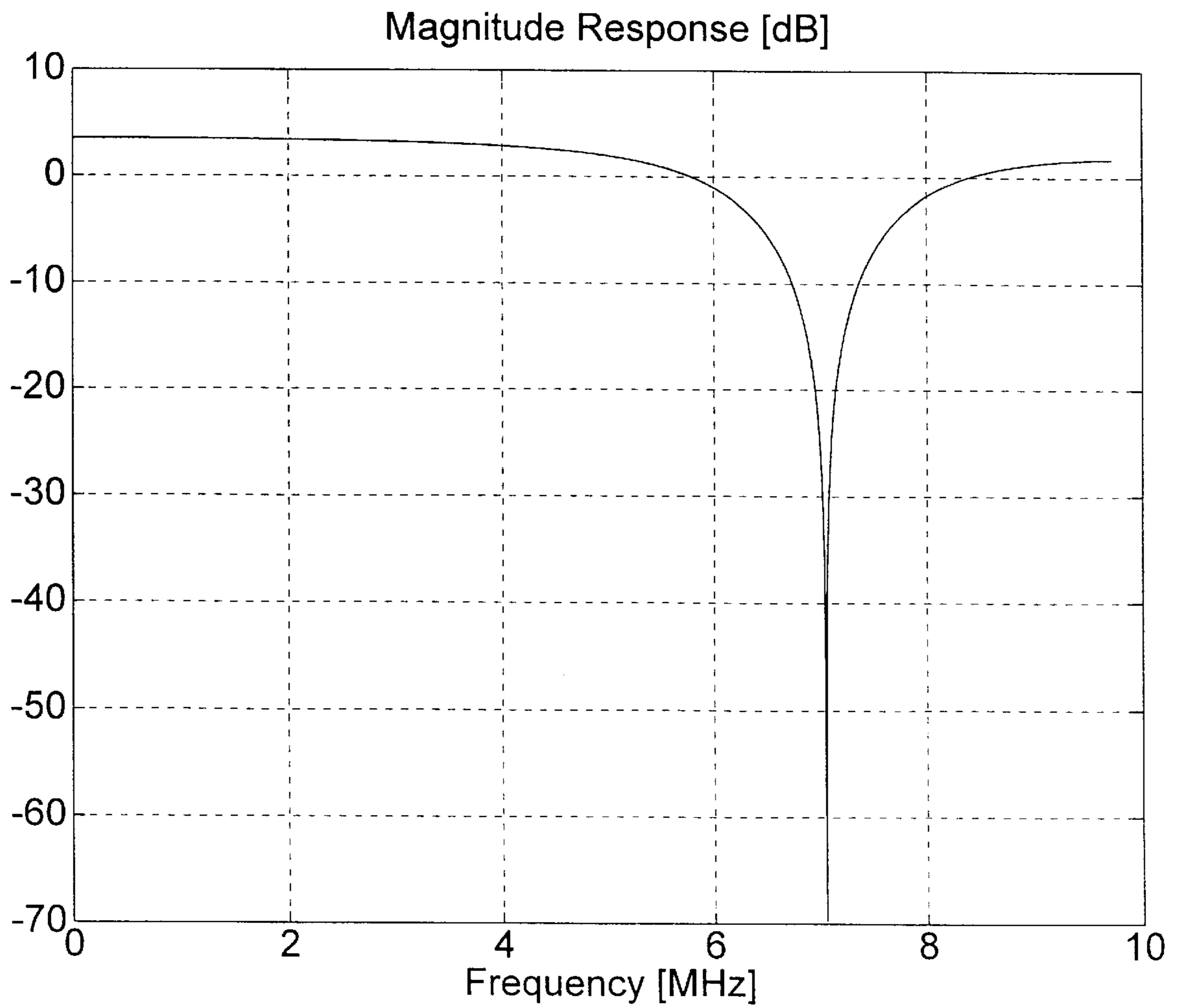


Fig. 7

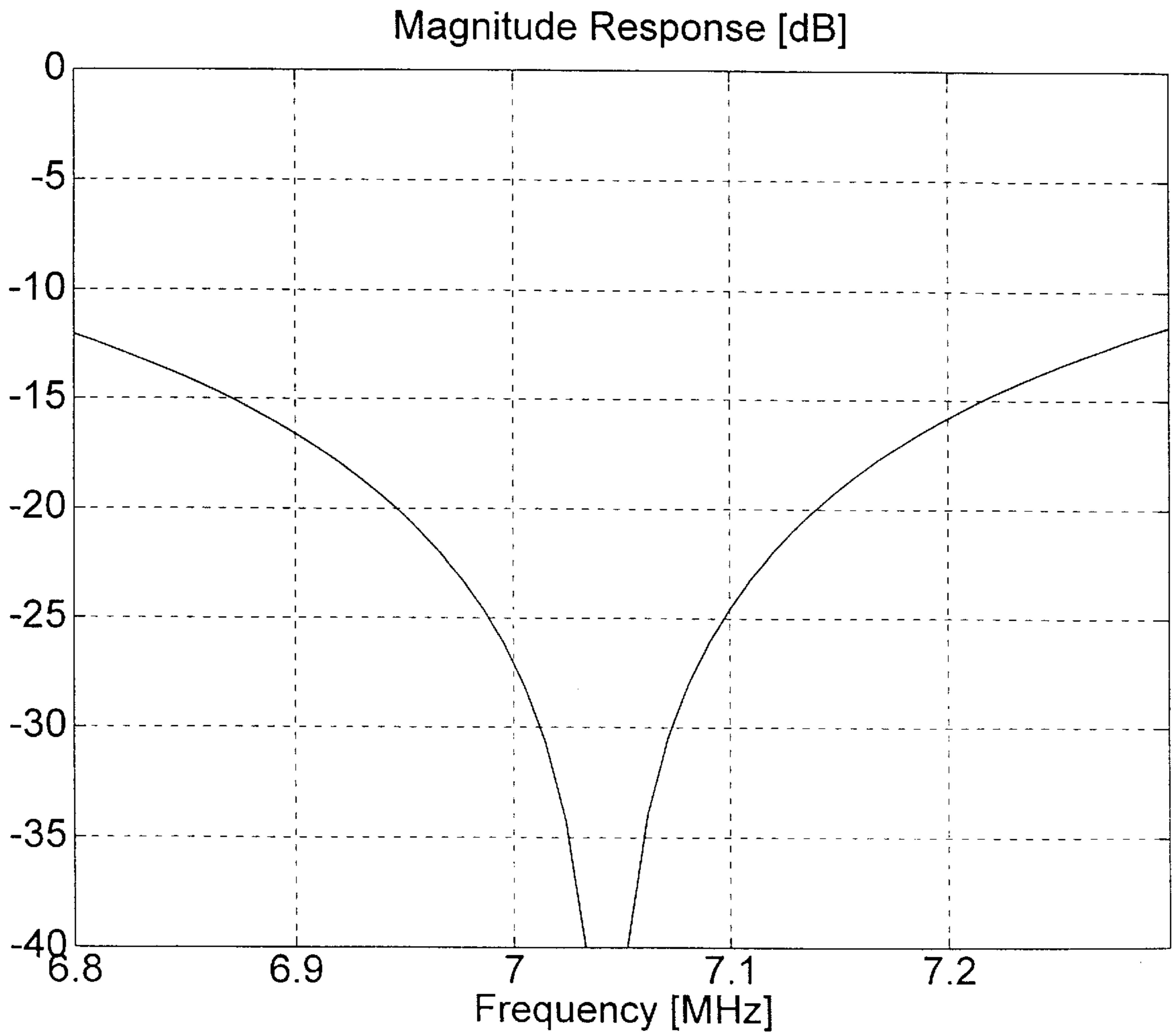


Fig. 8

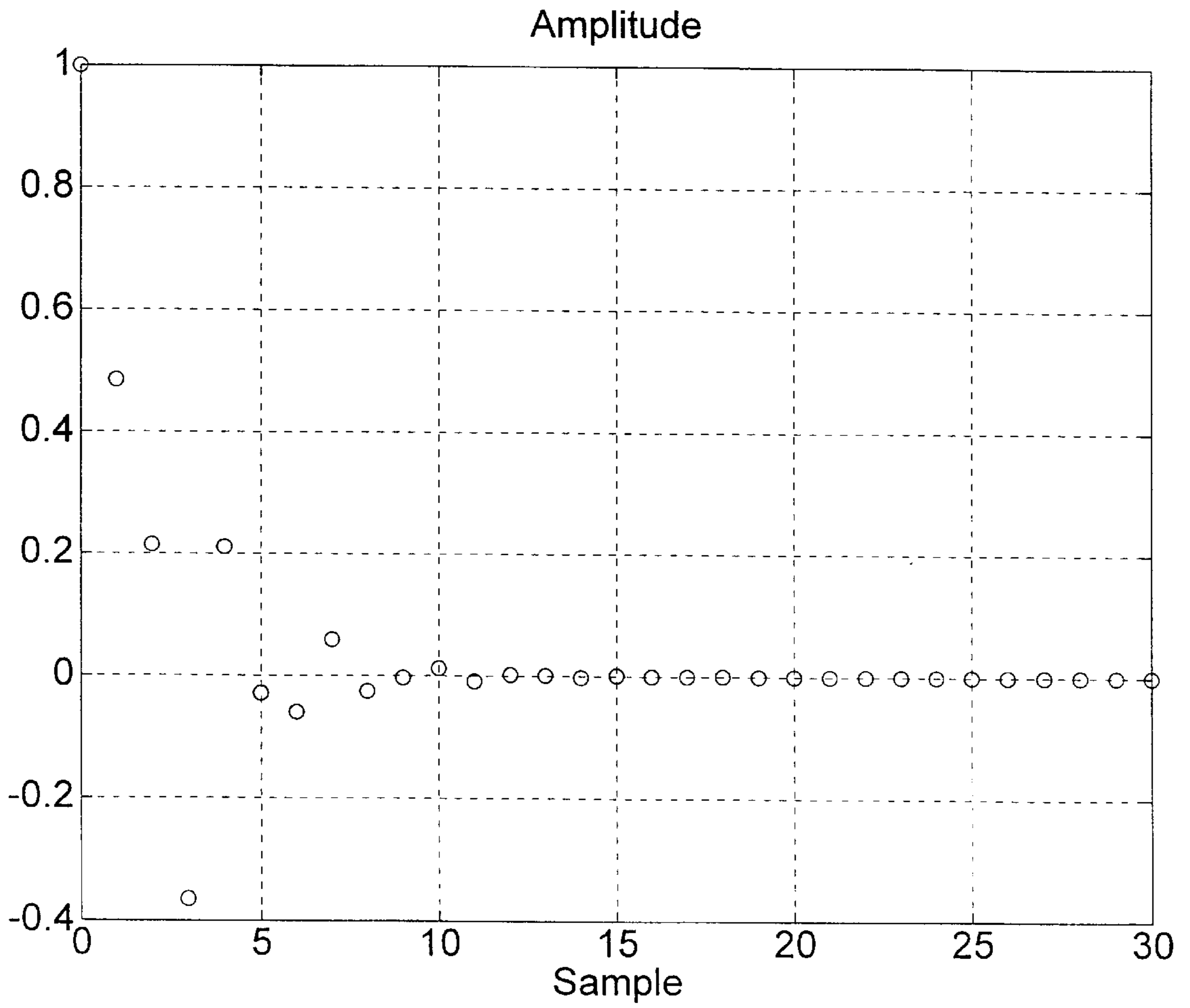


Fig. 9

## APPARATUS AND METHOD FOR TRANSMISSION OF HIGH SPEED DATA OVER COMMUNICATION CHANNELS

### FIELD OF THE INVENTION

The present invention relates to apparatus for wideband transmission of digital signals over telephone wires.

### BACKGROUND OF THE INVENTION

A CAP (carrierless amplitude and phase) modulator is described in U.S. Pat. No. 4,924,492 to Gitlin et al.

Use of adaptive notch filters having constrained poles and zeros to eliminate narrow-band or sine wave components with unknown frequencies from observed time series is described in Stoica, P. and Nehorai, A., "Performance analysis of an adaptive notch filter with constrained poles and zeros", IEEE Transactions on Acoustics, Speech and Signal Processing, 36(6), June 1988, and in Ng, T. S., "Some aspects of an adaptive digital notch filter with constrained poles and zeros", IEEE Transactions on Acoustics, Speech and Signal Processing, ASSP-35(2), February 1987.

The disclosures of all publications mentioned in the specification and of the publications cited therein are hereby incorporated by reference.

### SUMMARY OF THE INVENTION

The present invention seeks to provide improved modem apparatus.

The present invention relates to the transmission, at high bit rate, of digital signals over copper telephone wires. The pervading demand for high speed digital traffic requires the use of a high capacity physical medium as the backbone of a network. This is usually implemented via broadband optical fiber or coaxial cable. These cables interconnect the geographically distant central offices of the service providers (e.g. telephone companies) and also interconnect the central communications areas and curbside optical network units.

However, it is still extremely expensive to deploy fiber optic or coaxial cable to the individual premises of the customers. Therefore, the transmission over the local loop, typically the last 1-2 miles from an adjacent central office or from a curbside optical network unit to the customers' premises utilizes the existing infrastructure of twisted copper wires.

The local telephone line connecting a central office to the customers' premises is conventionally used for analog transmission of voice signals. DSL (digital subscriber line) technologies are used by telephone companies to provide high speed digital transmission over the local loop at reasonable cost. The use of copper local loops in conjunction with the backbone optical network enables broadband services to be provided to customers. There are various DSL technologies which vary in their transmission symmetries and transmission rates. The rate of transmission limits the service reach.

VDSL (very high bit rate digital subscriber line) is a type of DSL technology now being developed to carry high speed data for applications such as ATM (asynchronous transfer mode) and digital compressed video. VDSL technology increases the downstream data rate to about 13 to 52 Mb/s and the upstream data rate to about 1.6 to 6.5 Mb/s. These high rates require the use of frequency bands which are not used by other DSL technologies. A difficulty to be overcome is that the transmitted spectrum of a VDSL signal overlaps AM broadcasts and amateur radio transmissions causing RF

(radio frequency) egress and RF ingress. RF egress is the radiation from the VDSL transmission into amateur radio receivers. Field measurements have shown that the transmitted PSD (power spectral density) at amateur radio bands should be maintained at a low level to minimize interference, however the low level of PSD cannot be allowed to significantly compromise performance.

RF ingress is the induced power of the amateur radio transmission onto the VDSL receiver. The received interference may be stronger than the VDSL signal and state of the art digital receiving schemes may not be adequate to handle such a high level of noise. The frequency of amateur radio transmissions tends to drift, necessitating careful monitoring by the VDSL receiver.

There is thus provided, in accordance with a preferred embodiment of the present invention, apparatus for transmission of high speed data over communication channels, the apparatus including a modulator operative to modulate an outgoing stream of digital data, thereby to generate an outgoing signal, and a demodulator operative to demodulate an incoming signal, thereby to generate an incoming stream of digital data, wherein the modulator includes a band suppressor for suppressing portions of the outgoing signal which have specified frequencies.

Further in accordance with a preferred embodiment of the present invention, the specified frequencies include fixed predetermined frequencies and/or programmable frequencies.

Still further in accordance with a preferred embodiment of the present invention, the specified frequencies include a plurality of frequency bands.

Additionally in accordance with a preferred embodiment of the present invention, the band suppressor includes a plurality of band suppressor elements operative to suppress portions in the outgoing signal whose frequencies fall within the plurality of frequency bands respectively.

Still further in accordance with a preferred embodiment of the present invention, the modulator includes a single carrier modulator such as a quadrature amplitude modulator (QAM).

Also provided, in accordance with another preferred embodiment of the present invention, is apparatus for transmission of high speed data over communication channels, the apparatus including a modulator operative to modulate an outgoing stream of digital data, thereby to generate an outgoing signal, and a demodulator operative to demodulate an incoming signal, thereby to generate an incoming stream of digital data, wherein the demodulator includes a noise canceller including an adaptive IIR (infinite impulse response) digital filter operative to cancel narrow band interfering signals.

Further in accordance with a preferred embodiment of the present invention, the band suppressor includes at least one infinite impulse response (IIR) digital filter.

Still further in accordance with a preferred embodiment of the present invention, the demodulator also includes a noise canceller including a first adaptive digital filter receiving a first input signal and a second adaptive digital filter receiving a second input signal wherein the second input signal is received as an error signal by the first adaptive digital filter, wherein adaptation of the first adaptive digital filter is performed in accordance with the error signal and wherein an output of the noise canceller includes a combination of the outputs of the first and second adaptive digital filters.

Further in accordance with a preferred embodiment of the present invention, the demodulator includes a decision feed-

back equalizer and wherein the band suppressor is characterized in that any distortion of the incoming signal caused by the band suppressor is compensated substantially totally by the decision feedback equalizer.

Also provided, in accordance with another preferred embodiment of the present invention, is apparatus for transmission of high speed data over communication channels, the apparatus including a communication channel, a modulator operative to modulate an outgoing stream of digital data, thereby to supply an outgoing signal to the communication channel, and a demodulator operative to receive an incoming signal from the communication channel and to demodulate the incoming signal, thereby to generate an incoming stream of digital data, wherein the demodulator includes a band suppressor for suppressing portions of the incoming signal which have specified frequencies.

Further in accordance with a preferred embodiment of the present invention, the communication channel includes a two-wire telephone local loop.

Still further in accordance with a preferred embodiment of the present invention, the band suppressor includes an infinite impulse response digital filter.

Additionally in accordance with a preferred embodiment of the present invention, the infinite impulse response digital filter has an infinite impulse response which decays rapidly to a level close to zero.

Still further in accordance with a preferred embodiment of the present invention, the demodulator includes a decision feedback equalizer having a time span and wherein the infinite response digital filter has an infinite impulse response characterized in that the total energy of the infinite impulse response, summed over time, after the time span is small relative to the total energy of the infinite impulse response during the time span.

Further in accordance with a preferred embodiment of the present invention, the total energy of the infinite impulse response, summed over time, after the time span is less than  $\frac{1}{10}$ , or even less than  $\frac{1}{100}$ , or even less than  $\frac{1}{1000}$  of the total energy of the infinite impulse response during the time span.

Also provided, in accordance with another preferred embodiment of the present invention, is apparatus for transmission of high speed data over communication channels, the apparatus including a modulator operative to modulate an outgoing stream of digital data, thereby to generate an outgoing signal, and a demodulator operative to demodulate an incoming signal, thereby to generate an incoming stream of digital data, wherein the demodulator includes a noise canceller generating an output, the demodulator including a first adaptive digital filter receiving a first input signal and a second adaptive digital filter receiving a second input signal wherein the second input signal is received as an error signal by the first adaptive digital filter, wherein adaptation of the first adaptive digital filter is performed in accordance with the error signal and wherein the output of the noise canceller includes a combination of the outputs of the first and second adaptive digital filters.

Further in accordance with a preferred embodiment of the present invention, the first adaptive digital filter includes an adaptive infinite impulse response filter.

Still further in accordance with a preferred embodiment of the present invention, the demodulator includes a band suppressor for suppressing portions of the incoming signal which have specified frequencies.

Further in accordance with a preferred embodiment of the present invention, the demodulator also includes a noise

canceller including a first adaptive digital filter receiving a first input signal and a second adaptive digital filter receiving a second input signal wherein the second input signal is received as an error signal by the first adaptive digital filter, wherein adaptation of the first adaptive digital filter is performed in accordance with the error signal and wherein the output of the noise canceller includes a combination of the outputs of the first and second adaptive digital filters.

Still further in accordance with any of the above preferred embodiments of the present invention, the band suppressor includes a passband band suppressor and/or a baseband band suppressor.

Further in accordance with a preferred embodiment of the present invention, the first and/or second digital filters include a FIR filter.

In the present invention processing units are added to the conventional receiving scheme. These processing units are designed to decouple the egress and the ingress problems associated with conventional equalization machinery. The egress and the ingress issues are efficiently treated by these additional blocks.

#### BRIEF DESCRIPTION OF THE DRAWINGS

The present invention will be understood and appreciated from the following detailed description, taken in conjunction with the drawings in which:

FIG. 1A is a simplified block diagram of a modem device constructed and operative in accordance with a preferred embodiment of the present invention which includes a passband modulator band suppressor;

FIG. 1B is a simplified block diagram of a modem device constructed and operative in accordance with a preferred embodiment of the present invention which includes a baseband modulator band suppressor;

FIG. 2 is a simplified block diagram of a preferred implementation of one of the band suppressor devices of FIGS. 1A or 1B;

FIG. 3 is a simplified block diagram of a preferred passband implementation of an individual one of the band suppressor elements of FIG. 2, which implementation is suitable for the passband modulator of FIG. 1A;

FIG. 4 is a simplified block diagram of a preferred baseband implementation of an individual one of the band suppressor elements of FIG. 2, which implementation is suitable for the baseband modulator of FIG. 1B;

FIG. 5A is a simplified block diagram of a preferred implementation of the receiver of FIGS. 1A-1B, constructed and operative in accordance with a preferred embodiment of the present invention, which includes a passband demodulator band suppressor;

FIG. 5B is a simplified block diagram of a preferred implementation of the receiver of FIGS. 1A-1B, constructed and operative in accordance with a preferred embodiment of the present invention, which includes a baseband demodulator band suppressor;

FIG. 6 is a simplified block diagram of a preferred implementation of the noise canceller of FIGS. 5A-5B;

FIG. 7 is a graph of an example of a frequency response of the passband band suppressor of FIGS. 1A or 5A;

FIG. 8 is an enlarged portion of the graph of FIG. 7; and

FIG. 9 is a graph of the impulse response of a passband band suppressor having the frequency response illustrated in FIGS. 7-8.

#### DETAILED DESCRIPTION OF A PREFERRED EMBODIMENT

Reference is now made to FIG. 1A which is a simplified block diagram of a modem device constructed and operative

in accordance with a preferred embodiment of the present invention which includes a passband modulator band suppressor.

The modem illustrated herein is an example which is suitable for QAM (quadrature amplitude modulation) applications. However it is appreciated that the passband band suppressor of FIGS. 2-4 and the noise canceller of FIGS. 5A-6 are suitable, more generally, for any type of single carrier modulation application and for CAP (carrier-less amplitude phase) modulators.

The apparatus of FIG. 1A includes a hybrid line coupling unit 10 linking a transmitter 20 and a receiver 30 to a 2-wire telephone local loop 40. The hybrid line coupling unit 10 conveys a transmitted signal to local loop 40 and couples a received signal to the receiver 30. Both signals use the same pair of wires for full duplex communications. The hybrid line coupling unit 10 connects the 2-wire telephone local loop 40 to two wire pairs, associated respectively with a transmitter amplifier 50 and a receiver filter 60 (FIGS. 5A-5B), which are used for transmission and reception respectively. A description of how to construct one example of a suitable hybrid line coupling unit 10 is provided in Chen, W. Y. et al, "High bit rate digital subscriber line echo cancellation", IEEE Journal on Selected Areas in Communications, 9(6), August 1991.

The transmitter 20 preferably includes the following elements:

A conventional bit-to-symbol mapper 70 is provided, via which data is routed to a modulator shaping filter 80 which typically comprises two finite impulse response filters (FIR) including an in-phase filter and a quadrature filter. The filter 80 independently filters two streams of samples, namely the in-phase portion of the incoming data and the quadrature portion of the incoming data, thereby generating two baseband signal streams. The filters may, for example, comprise Nyquist pulses of the type described in Chapter 4 of E. A. Lee and D. G. Messerschmitt, *Digital Communication*, Kluwer Academic Publishers, Boston, 1988.

An up-converter 90 converts the two baseband signal streams into a single passband signal, typically by multiplying the two streams respectively by two sinusoids having a  $\pi/2$  phase difference and subtracting one from the other.

An optional passband modulator band suppressor 100 receives the output of the up-converter 90. The band suppressor 100 preferably includes a sequence of one or more band suppressor elements 110, such as three band suppressor elements, as shown in FIG. 2. Each band suppressor element is operative to suppress portions of the outgoing signal which fall within a particular frequency band.

A preferred embodiment of an individual band suppressor element 110 is described in detail below with reference to FIG. 3. Each band suppressor element 110 creates a notch having a distinct  $(f_n, W)$  pair.  $f_n$ , which is a function of  $c$ , a multiplication factor in FIG. 3, is the center frequency of the band suppressor element and  $W$ , which is a function of  $r$ , another multiplication factor in FIG. 3, is bandwidth. A sequence of band suppressor elements 110 such as that shown in FIG. 2 implements a multi-notched mask which provides a low level of transmitted PSD (power spectral density) in certain spectral bands such as amateur radio bands. More generally, a low level of transmitted PSD is preferably provided in spectral bands which are shared by other transceivers which are vulnerable to a DSL transmission.

FIG. 3 is a simplified block diagram of a preferred implementation of an individual one of the band suppressor

elements 110 of FIG. 2. Each of the elements of FIG. 3 may have parameters  $c$  and/or  $r$  which are either programmable or not programmable. In the illustrated embodiment, the  $r$  parameter is shown to be programmable and the  $c$  parameter is shown to be predetermined however it is appreciated that this is not intended to be limiting.

The difference equation of the filter of FIG. 3 is given by:

$$y(l)=x(l)+cx(l-1)+x(l-2)-rcy(l-1)-r^2 y(l-2).$$

where:

$x(l)$ =the input to the filter of FIG. 3 at time instant  $l$ ;  
 $y(l)$ =the output from the filter of FIG. 3 at time instant  $l$ ;  
 $c$  is a function of the center frequency  $f_n$  of the notched band:

$$c=-2 \cos(2\pi f_n/f_s);$$

and

$f_s$  is the sample rate.

$r$  is a parameter in the range  $[0,1]$  and is typically only slightly less than 1.  $r$  determines the bandwidth of the notch filter in that the notch becomes narrow as  $r$  approaches 1.

A similar notch filter is described in the above-referenced Stoica and Nehorai publication.

A particular feature of a preferred embodiment of the present invention is that if a first modem transmits to a second modem, the band suppressor 100 in the first modem has an infinite impulse response whose total energy summed over time, after the time span of the decision feedback equalizer 230 (FIGS. 5A-5B) of the second modem, is small relative to the total energy of the infinite impulse response.

Referring again to FIG. 1A, the output of the band suppressor is received by a digital to analog converter 120. In FIG. 1B, in contrast, the D/A converter 120 receives the output of up-converter 90. D/A converter 120 may, for example, comprise an AD9721 D/A converter, commercially available from Analog Devices.

Transmitter filter 130 rejects high frequency images and spurious signals present in the output of D/A converter 120. The configuration of transmitter filter 130 depends on the transmitted frequency band and the out-of-band spectral purity requirements of the particular application.

Transmitter amplifier 50 amplifies the signal to the desired level for transmission.

The hybrid line coupling 10 couples the outgoing signal to the physical medium, i.e. local loop 40.

FIG. 1B is a simplified block diagram of a modem device constructed and operative in accordance with a preferred embodiment of the present invention which includes a baseband modulator band suppressor 134. The baseband modulator band suppressor 134 may, like passband modulator band suppressor 100, include a sequence of one or more suppressor elements 110 as shown in FIG. 2, one of which is illustrated in FIG. 4. Since the baseband modulator band suppressor 134 precedes the up-converter 90, it processes complex valued samples.

The difference equation of the filter of FIG. 4 is:

$$y(m)=x(m)+Cx(m-1)-R C y(m-1).$$

where:

$x(m)$ =the input to the filter of FIG. 4 at time instant  $m$ ;

$y(m)$  = the output from the filter of FIG. 4 at time instant  $m$ ;

$C$  is a function of the center frequency  $f_n$  of the notched band:

$$C = -\exp(2\pi f_n / f_s);$$

$f_s$  is the processing rate;

and wherein  $x(m)$ ,  $y(m)$  and  $C$  are all complex valued quantities.

$R$  is a parameter in the range (0,1) and is typically only slightly less than 1.  $R$  determines the bandwidth of the notch filter in that the notch becomes narrow as  $R$  approaches 1.

Two embodiments of the receiver **30** of FIGS. 1A–1B are now described in detail with reference to FIGS. 5A–5B:

The hybrid line coupling **10** of FIGS. 1A–1B is operative to receive a far end signal via two-wire cable **40** and to transfer the received signal to the receiver filter **60** of the receiver **30**.

In the analog domain, the received signal is processed by receiver filter **60** and receiver amplifier **140** which typically comprises an automatic gain control amplifier. The receiver filter **60** rejects the received power at frequencies falling outside of the system's transmission band. The resulting signal passes to amplifier **140** whose gain is typically adapted in accordance with the level of the received signal. The amplifier **140** includes a control loop which maintains a constant power signal at the input to an A/D converter **150**. This control is enabled only at the start-up stage. At the end of this stage, the automatic gain control is preferably terminated since continual operation of the circuit may prevent proper cancellation of varying power interferences. Typically, the communication characteristics of channel **40** vary slowly over time and compensation for this effect is preferably provided by a decision feedback equalizer unit **230**, described in detail below.

The analog output signal of analog amplifier **140** is converted to quantized samples by the A/D converter **150**. The A/D converter **150** may, for example, comprise an AD9040AD/A converter, commercially available from Analog Devices. The level of the input signal to the D/A converter, when free of narrowband interferences, is typically set to occupy less than the full precision scale of the A/D so as to leave a few MSBs (most significant bits), such as 1–3 bits, for the digital representation of high level interfering signals.

The sampling epochs of the A/D are determined by a VCO (voltage controlled oscillator) **160**. A clock recovery circuit **170** controls the oscillator **160** and thereby recovers a clock at an integer multiple of the transmitted symbol rate. The VCO **160** in conjunction with the clock recovery circuit **170** generate a predetermined number of clock pulses, such as 2–4 clock pulses, every symbol period. The integer multiple of the transmitted symbol rate is typically equal to the processing rate of the demodulator, expressed in samples/symbol.

Several techniques for timing recovery are described in Chapter 15 of the above-referenced Lee and Messerschmitt publication and in Chapter 6 of R. D. Gitlin et al, *Data Communications Principles*, Plenum Press, New York, 1992.

The embodiment of FIG. 5A includes a passband demodulator band suppressor **180** whereas the embodiment of FIG. 5B includes a baseband demodulator band suppressor **234** as described in detail below. The receiver band

suppressor **180** may be similar to the band suppressor of FIGS. 2–3 except that it may reject different frequency ranges. The receiver band suppressor **180** provides a predetermined rejection of notched bands which are selected based on the application. The output signal of the band suppressor **180** is routed to a down converter **190** which converts the passband real-valued samples to baseband complex valued samples. This is performed by multiplying the real valued samples by  $\exp\{-j 2\pi f_c n/f_s\}$ , where:

$$j = (-1)^{1/2};$$

$f_c$  = the center frequency of the passband signal;

$f_s$  = the processing rate; and

$n$  = the serial number of a current sample.

As illustrated in FIG. 5A, the subunits which follow the A/D converter **150** form an adaptive scheme. This scheme includes a FFE (feedforward equalizer) **200**, a DFE (Decision feedback equalizer) **230** and a narrow band noise canceller **210**.

The output of the down converter **190** is fed into the FFE **200**. The decisions of a slicer **220** are routed to the noise canceller **210** and to the DFE **230**. The sum of the output samples of the FFE **200** and the DFE **230** is fed into the noise canceller **210**. The error signal generated by the slicer **220** is used to adapt all the three filters. Additionally the error signal is processed by the noise canceller **210** to generate the output samples.

The FFE **200** partially compensates for phase and amplitude distortion of the channel **40**. The FFE **200** preferably operates as a fractionally spaced equalizer in that it processes the data at a higher rate than the symbol rate. An advantage of fractionally spaced equalization operation is that aliasing of the received signal is avoided.

The DFE **230** typically comprises a conventional adaptive filter that unravels ISI (inter-symbol interference) caused by the channel with a reduced noise enhancement relative to the FFE **200**. Both filters **200** and **230** are adapted using an error signal generated by a slicer **220**. The error signal indicates the difference between the signal entering the slicer and the output signal of the system, which the slicer **220** generates and feeds back to the DFE **230** as well as out of the system.

Additional design considerations for filters **200** and **230** are described in Chapter 9 of the above-referenced Lee and Messerschmitt publication, in Chapter 8 of the above-referenced Gitlin publication and in S. U. Qureshi, "Adaptive equalization", Proceedings of the IEEE, Vol. 73(9), September 1985, pp. 1349–1387.

Reference is now made to FIG. 5B which is a simplified block diagram of a preferred implementation of the receiver of FIGS. 1A–1B, constructed and operative in accordance with a preferred embodiment of the present invention, which includes a baseband demodulator band suppressor. The apparatus of FIG. 5B is generally similar to the apparatus of FIG. 5A except for the following differences:

The passband band suppressor **180**, preceding down-converter **190** in FIG. 5A, is replaced by a baseband band suppressor **234** which follows down-converter **190**. As described above, the passband band suppressor **180**, as illustrated in FIG. 2, may include one or more of the passband band suppressor elements of FIG. 3. In contrast, the baseband band suppressor **234**, as illustrated in FIG. 2, may include one or more of the baseband band suppressor elements of FIG. 4.

FIG. 6 is a simplified block diagram of a preferred implementation of the noise canceller **210** of FIGS. 5A–5B. Preferably, the noise canceller **210** includes a pair of adaptive filters **250** and **260**, arranged as shown, which together provide strong and rapid rejection of narrow band interferences.

The apparatus of FIGS. 5A-6 decouples ISI mitigation from RFI (radio frequency interference) cancellation by achieving the two effects with two separate mechanisms. The filters 200 and 230 of FIGS. 5A-5B mitigate ISI and the adaptive filters 250 and 260 of FIG. 6 cancel radio frequency interfering signals. The narrow band noise canceller of FIG. 6 receives the slicer decisions and a sum  $X_p$  of the outputs of FFE 200 and DFE 230. A non-zero difference between these two samples is indicative of the existence of a narrow band interference signal. This difference is fed to the first adaptive filter 250.

If narrow band interference occurs during the start-up period of the apparatus shown and described herein, the apparatus is typically restarted. Once the start-up period has passed and the DFE and FFE have already converged, then narrow band interference of intermittent transmission, if present, is nulled by the narrow band noise canceller of FIG. 6, and the DFE and FFE, consequently, need not adapt to a different solution. The adaptive filter 250 of narrow band noise canceller of FIG. 6 generates estimates of the instantaneous level and frequency of RFI at the point of input to the apparatus of FIG. 6. Therefore, the RFI interference can be cancelled out.

The separate treatment of RFI and ISI by the noise canceller of FIG. 6, and elements 200 and 230, respectively, ensures both rapid tracking of the frequency-agile interfering signals and strong rejection.

Further acceleration of the cancellation process is achieved by providing two filters in the structure of FIG. 6. The first filter 250 comprises an FIR filter or an IIR (infinite impulse response) filter which has very low noise enhancement. The second filter 260 typically comprises a FIR structure which has a higher level of noise enhancement and a very short convergence time.

When a transmission of a narrow band interfering signal occurs, the second filter 260 provides the initial cancellation of the signal, due to its very short convergence time. Subsequently, the first filter develops a notch at the interference frequency and the notch of the second filter vanishes. Therefore, the steady state performance in the presence of narrow band interfering signals degrades only slightly relative to performance in a normal situation in which narrow band interference is absent.

The narrow band noise canceller of FIG. 6 processes substantially only the noise portion of the incoming signal received by the system from two-wire channel 40 and therefore, unlike conventional standard notch filters, does not generate additional ISI.

Preferably, since ISI is dealt with by filters 200 and 230 of FIGS. 5A-5B, the narrow band noise canceller of FIG. 6 is configured to deal with only RFI, but without increasing the amount of ISI.

Preferred relationships between the input and output quantities of FIG. 6 are now described, using the following notation:

$X_p$ =input sample of the narrow band noise canceller of FIG. 6 at time instant p;

$Y_p$ =output sample of the narrow band noise canceller of FIG. 6 at time instant p;

$A_p$ ="slicer decision" output of slicer 220 at time instant p, normalized to the level of the corresponding input signal to the slicer.

$E_p$ =the corresponding error sample generated by the slicer at the same time instant p, defined as:

$$E_p = Y_p - A_p$$

$F_1, F_2, \dots, F_k$ =adaptive complex valued coefficients of the first adaptive filter 250 which is a constrained IIR structure, typical constraints being defined below

$S$ =real valued parameter of the first filter 250, in the [0,1] range, typically a value close to 1 such as 0.875.

$G_1, G_2, \dots, G_q$ =adaptive coefficients of the second adaptive filter 260 which typically comprises an adaptive FIR.

$U_p$ =the input to the first filter at time instant p, namely:

$$U_p = X_p - A_p$$

$V_p$ =the output from the first filter at time instant p, namely:

$$V_p = \sum [-S^i (U_{p-i} + V_{p-i}) + U_{p-i}] F_i$$

$$i=1, 2, \dots, k$$

$Z_p$ =an intermediate sample, as shown in FIG. 6, namely

$$Z_p = V_p + X_p$$

$W_p$ =the input signal to the second filter 260, namely

$$W_p = Z_p - A_p$$

$H_p$ =the output from the second filter 260 at time instant p.

The output of the apparatus of FIG. 6,  $Y_p$ , as shown in FIG. 6, is:

$$Y_p = X_p + V_p + \sum G_i W_{p-i}$$

$$i=1, 2, \dots, q$$

The second filter 260 is adapted according to a suitable criterion such as the LMS (least mean squares) criterion, using the error samples generated by the slicer. Conventional LMS based adaptation processes are described in Chapter 9 of the above-referenced Lee and Messerschmitt publication, in Chapter 8 of the above-referenced Gitlin publication and in the above referenced Qureshi article.

Adaptation of the first filter may be a conventional adaptation process applicable to IIR filters as described in Regalia, P. A., *Adaptive IIR filtering in signal processing and control*, Marcel Dekker, New York, 1995.

Alternatively, a simpler adaptation rule, conventionally used for FIR filters, may be employed, namely:

$$F_d \leftarrow F_d - \mu W_g (U_{g-d})^*$$

where:

$\mu$ =adaptation constant of the first filter

$F_d$ =the d-th coefficient of the first filter (d=1,2, ..., k)

\*=conjugate operator

g=time instant

It is appreciated that optionally, the second filter 260 may be omitted along with the summer 262, the summer 264 and the delay 266. In this embodiment, the first filter is typically an IIR filter.

Referring back to FIGS. 5A-5B, the slicer 220 is now described. The slicer generates estimates, also termed herein "slicer decisions", of the sequence of symbols generated by bit to symbol mapper 70 from the transmitted data arriving from a remote transmitter. The estimate at any decision instant typically is taken to be the nearest symbol to the



slicer input. The slicer also generates error samples,  $E_p$ , which are the differences between the input samples and the decisions made therefrom.

A suitable slicer is described in Chapter 4 of the above-referenced Lee and Messerschmitt publication.

FIG. 7 is a graph of an example of a frequency response of the passband modulator band suppressor of FIG. 1A for a notch whose center frequency is 7.05 MHz. In the example of FIG. 7, the processing rate is 19.44 MHz, the symbol rate is 6.48 MBaud. As best seen in FIG. 8, this filter provides more than 20 dB of rejection of the amateur frequency band 7–7.1 MHz if, for example,  $r=3/4$  and  $c=83/64$ .

FIG. 9 is a graph of the impulse response of a passband modulator band suppressor having the frequency response illustrated in FIGS. 7–8. FIG. 9 shows that after 15 samples (equivalent to 5 symbol periods), the absolute value of the impulse response is very small, typically smaller than 0.01. After 23 samples (equivalent to about 8 symbol periods), the absolute value of the impulse response is smaller than 0.001.

It is appreciated that, according to alternative embodiments of the present invention, a modem may include one, some or all of the following elements: the passband modulator band suppressor **100** of FIG. 1A, the baseband modulator band suppressor **134** of FIG. 2, the passband demodulator band suppressor **180** of FIG. 5A, the baseband demodulator band suppressor **234** of FIG. 5B and the noise canceller of FIG. 6. A modem including any subset of the above elements is within the scope of the present invention.

The present invention is disclosed using a QAM modulation. However, the modules in the present invention which are to suppress narrow band interferers and those units which are used to reduce the transmit PSD in given frequency bands are utilized in conjunction with any single carrier modulation technique. This method includes other two-dimensional schemes and also higher dimensional schemes. The band suppressors and the narrow band noise canceller may also be implemented in baseband modulation and carrierless modulation schemes.

The present invention is disclosed in the context of the VDSL transmission technique which operates at downstream bit rates of at least 6 Mb/s. Additionally the present invention may also be used at lower bit rates.

It is appreciated that the software components of the present invention may, if desired, be implemented in ROM (read-only memory) form. The software components may, generally, be implemented in hardware, if desired, using conventional techniques.

It is appreciated that various features of the invention which are, for clarity, described in the contexts of separate embodiments may also be provided in combination in a single embodiment. Conversely, various features of the invention which are, for brevity, described in the context of a single embodiment may also be provided separately or in any suitable subcombination.

It will be appreciated by persons skilled in the art that the present invention is not limited to what has been particularly shown and described hereinabove. Rather, the scope of the present invention is defined only by the claims that follow:

I claim:

1. Apparatus for reception of high speed data from a noisy communication channel, the apparatus comprising:

a demodulator at one end of the noisy communication channel which receives and demodulates a primary input comprising signal and noise, the demodulator comprising:

a decision unit generating a decision output  $A_p$  from the primary input;

a noise canceller receiving a noise canceller input signal  $X_p$  and generating a noise canceller output signal  $Y_p$ , the noise canceller including:

a first adaptive digital filter receiving a first filter input signal  $U_p$  and generating a first filter output signal  $V_p$ , and a second adaptive digital filter receiving a second filter input signal  $W_p$  and generating a second filter output signal  $H_p$ ;

a first adder for combining first and second adder inputs  $X_p$  and  $A_p$  into a first adder output  $U_p$ , the first adder including a first delay unit delaying said first adder input, said first adder input comprising said noise canceller input signal  $X_p$ , said second adder input comprising decision output  $A_p$ , said first adder output comprising the first filter input signal  $U_p$  to the first adaptive digital filter;

a second adder for combining third and fourth adder inputs  $X_p$  and  $V_p$  into a second adder output  $Z_p$ , said third adder input comprising the noise canceller input signal  $X_p$ , said fourth adder input comprising the first filter output signal  $V_p$ ;

a third adder for combining fifth and sixth adder inputs  $Z_p$  and  $A_p$  into a third adder output  $W_p$ , the third adder including a second delay unit delaying said fifth adder input, said fifth adder input comprising the second adder output  $Z_p$ , said sixth adder input comprising the decision output  $A_p$ , and wherein the third adder output  $W_p$  comprises the second filter input signal;

a fourth adder for combining seventh and eighth adder inputs  $Z_p$  and  $H_p$  into a fourth adder output  $Y_p$ , said seventh adder input comprising the second adder output  $Z_p$ , said eighth adder input comprising the second filter output signal  $H_p$ , and

wherein the noise canceller output signal  $Y_p$  is a function of the sum of the first and second filter output signals  $V_p$  and  $H_p$ .

2. Apparatus according to claim 1 wherein said first adaptive digital filter comprises an adaptive infinite impulse response filter.

3. Apparatus according to claim 1 wherein said second digital filter comprises a FIR filter.

4. Apparatus according to claim 1 wherein said first digital filter comprises a FIR filter.

5. Apparatus according to claim 1 and also comprising a modulator operative to modulate an outgoing stream of digital data, thereby to generate an outgoing signal which is transmitted over the noisy communication channel to generate said incoming signal.

6. Apparatus according to claim 1 wherein the second filter input signal  $W_p$  is fed to the first filter as an error signal and wherein coefficients of the first filter are modified responsive to the value of the error signal.

7. Apparatus for reception of high speed data from a noisy communication channel, the apparatus comprising:

a demodulator at one end of the noisy communication channel which receives and demodulates a primary input comprising signal and noise, the demodulator comprising:

a decision unit generating a decision output  $A_p$  from the primary input;

a noise canceller receiving a noise canceller input signal  $X_p$  and generating a noise canceller output signal  $Z_p$ , the noise canceller including:

an adaptive IIR digital filter receiving a filter input signal  $U_p$  and generating a filter output signal  $V_p$ ;

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a first adder for combining first and second adder inputs  $X_p$  and  $A_p$  into a first adder output  $U_p$ , the first adder including a first delay unit delaying said first adder input, said first adder input comprising said noise canceller input signal  $X_p$ , said second adder input comprising decision output  $A_p$ , said first adder output comprising the filter input signal  $U_p$  to the adaptive digital filter; and

a second adder for combining third and fourth adder inputs  $X_p$  and  $V_p$  into a second adder output  $Z_p$ , said third adder input comprising the noise canceller input signal  $X_p$ , said fourth adder input comprising the filter output signal  $V_p$ ,

wherein the output of the noise canceller comprises a function of the filter output signal  $V_p$ .

8. Apparatus according to claim 7 and also comprising a modulator operative to modulate an outgoing stream of digital data, thereby to generate an outgoing signal which is transmitted over the noisy communication channel to generate said incoming signal.

9. A method for reception of high speed data from a noisy communication channel, the method comprising:

employing a demodulator at one end of the noisy communication channel to receive and demodulate a primary input comprising signal and noise, the employing step comprising:

generating a decision output  $A_p$  from the primary input; receiving a noise canceller input signal  $X_p$  and generating a noise canceller output signal  $Y_p$ , including:

providing a first filter input signal  $U_p$  and generating a first filter output signal  $V_p$ , and providing a second filter input signal  $W_p$  and generating a second filter output signal  $H_p$ ;

combining first and second adder inputs  $X_p$  and  $A_p$  into a first adder output  $U_p$ , including delaying said first adder input, said first adder input comprising said noise canceller input signal  $X_p$ , said second adder input comprising decision output  $A_p$ , said first adder output comprising the first filter input signal  $U_p$  to a first adaptive digital filter;

combining third and fourth adder inputs  $X_p$  and  $V_p$  into a second adder output  $Z_p$ , said third adder input comprising the noise canceller input signal  $X_p$ , said fourth adder input comprising the first filter output signal  $V_p$ ;

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combining fifth and sixth adder inputs  $Z_p$  and  $A_p$  into a third adder output  $W_p$ , including delaying said fifth adder input, said fifth adder input comprising the second adder output  $Z_p$ , said sixth adder input comprising the decision output  $A_p$ , and wherein the third adder output  $W_p$  comprises the second filter input signal;

combining seventh and eighth adder inputs  $Z_p$  and  $H_p$  into a fourth adder output  $Y_p$ , said seventh adder input comprising the second adder output  $Z_p$ , said eighth adder input comprising the second filter output signal  $H_p$ , and

wherein the noise canceller output signal  $Y_p$  is a function of the sum of the first and second filter output signals  $V_p$  and  $H_p$ .

10. A method for reception of high speed data from a noisy communication channel, the method comprising:

employing a demodulator at one end of the noisy communication channel to receive and demodulate a primary input comprising signal and noise, the employing step comprising:

generating a decision output  $A_p$  from the primary input; receiving a noise canceller input signal  $X_p$  and generating a noise canceller output signal  $Z_p$ , including: providing a filter input signal  $U_p$  and generating a filter output signal  $V_p$ ;

combining first and second adder inputs  $X_p$  and  $A_p$  into a first adder output  $U_p$ , including delaying said first adder input, said first adder input comprising said noise canceller input signal  $X_p$ , said second adder input comprising decision output  $A_p$ , said first adder output comprising the filter input signal  $U_p$  to an adaptive digital filter; and

combining third and fourth adder inputs  $x_p$  and  $V_p$  into a second adder output  $Z_p$ , said third adder input comprising the noise canceller input signal  $X_p$ , said fourth adder input comprising the filter output signal  $V_p$ ,

wherein the output of the noise canceller comprises a function of the filter output signal  $V_p$ .

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